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⑤ Int. Cl.4: H03L 7/08 , H03L 7/22

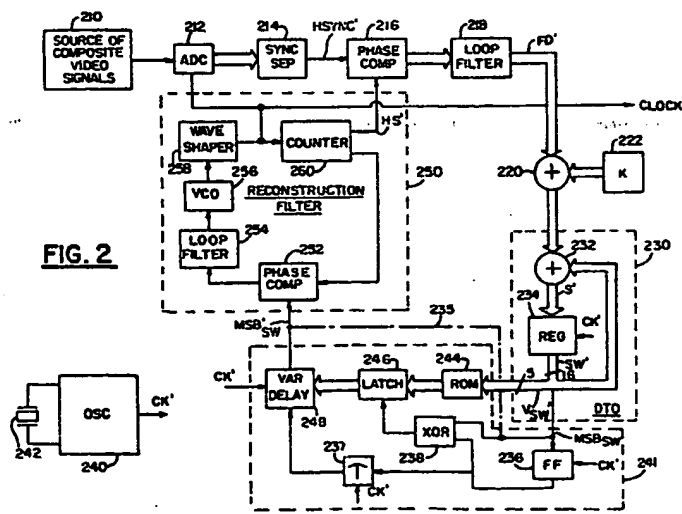
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DIGITALLY CONTROLLED PHASE LOCKED LOOP SYSTEM

The present invention relates to a hybrid phase locked loop system which includes analog and digital circuitry.

Recent developments in the field of digital television signal processing have generated interest in digitally controlled phase locked loop (PLL) systems. These systems have been shown to be useful in generating clock signals that are phase-locked to the horizontal line scanning signals used in a multi-standard (i.e. PAL or NTSC) television receiver. A system of this type is described in an article by A.H.H.J. Nillesen entitled "Line-locked Digital Colour Decoding", Electronic Components and Applications, Vol. 7, No. 4, 1988, pp. 239-245.

The PLL system described in the Nillesen article is a dual phase locked loop. An outer digital loop which includes a phase comparator, loop filter, adder, digital value source, DTO, ROM and DAC controls an inner analog PLL used as a "reconstruction filter". In this dual loop system, the digital loop operates at a relatively low frequency, f_{sw} , which is less than the Nyquist limit frequency for the clock signal which clocks the DTO. The signal produced by the digital PLL is multiplied in frequency by the analog PLL in the reconstruction filter to produce a sampling clock signal (CLOCK) for use in digitizing composite video signals in an ADC (analog to digital converter). The signal CLOCK is divided in frequency by a counter to produce a signal which is applied to an analog phase comparator of the inner PLL.

In the Nillesen system, the digital values produced by the ROM are converted into an analog signal by the DAC for application to the analog phase comparator of the reconstruction filter. The DAC and analog phase comparator of the Nillesen system are intended to be realized as a portion of an integrated circuit. Because these circuits are analog, they are very sensitive to variations in design and manufacturing parameters and may be subject to degradation in performance as the integrated circuit ages. It is herein recognized as desirable if digital circuitry, which is less sensitive to parametric variations and to aging, could be substituted for the DAC and analog phase comparator.

The present invention is embodied in a partly digital, partly analog phase locked loop system. The system includes an analog oscillator which is responsive to a control signal for generating an analog oscillatory signal. The system further includes a digital oscillator which generates a multi-bit digital oscillatory signal having a frequency which approximates a predetermined frequency. A phase comparator is responsive to the output signal of the analog oscillator and to one of the bits of the multi-bit digital oscillatory signal to develop the control signal applied to the analog oscillator.

FIGURE 1 is a block diagram of a partly analog, partly digital (prior art) phase locked loop system.

FIGURE 2 is a block diagram of a partly analog, partly digital phase locked loop system incorporating the present invention.

FIGURES 3A, 3B, 3C and 3D are waveform diagrams of signal amplitude versus time showing signals that are useful for explaining the operation of the circuitry shown in FIGURE 2.

In the drawings, broad arrows represent busses for conveying multiple-bit parallel digital signals and line arrows represent connections for conveying analog signals or single-bit digital signals. Depending on the processing speed of the devices, compensating delays may be required in certain of the signal paths. One skilled in the art of digital and analog signal processing circuit design would know where such delays would be needed in a particular system.

The line-locked clock system described in the Nillesen article is shown in FIGURE 1, labeled "Prior Art". The system shown in FIGURE 1 is a compound phase-locked loop. Generally, an analog PLL 50 is used as a "reconstruction filter" in a digital PLL which also includes a phase comparator 16, a loop filter 18 and a digital oscillator, also known as a discrete time oscillator (DTO) 30. The DTO 30 is clocked by a crystal controlled clock signal generated by an analog oscillator 40.

The following is a more detailed description of the circuitry shown in FIGURE 1. A source of composite video signals 10, which includes a conventional television tuner, IF amplifier and video detector, provides composite video signals to an analog to digital converter (ADC) 12. The ADC 12 digitizes the composite video signals at instants determined by a sampling clock signal, CLOCK, provided by the wave shaping circuitry 58 of the reconstruction filter 50. The digitized composite video signal developed by the ADC 12 is applied to synchronization signal separating circuitry 14. The circuitry 14 extracts the horizontal line synchronizing signal components from the digitized composite video signal and provides these extracted components as a signal HSYNC. The signal HSYNC is applied to one input port of the phase comparator 16. Another input port of the phase comparator 16 is coupled to receive a signal, HS, provided by a counter 60 of the reconstruction filter 50. The counter 60 frequency divides the signal CLOCK to generate the signal HS having a frequency, F_L , which approximates the frequency of the horizontal line synchronization signal

HSYNC.

The phase comparator 16 generates a signal proportional to the difference in phase between the signals HS and HSYNC. The signal provided by the phase comparator 16 is applied to the loop filter 18. The filter 18 integrates the phase difference signal to produce a signal, FD, which is indicative of the difference in frequency between the signals HS and HSYNC. The signal provided by the loop filter 18 is applied to one input port of an adder 20. The other input port of the adder 20 is coupled to a digital value source 22 which applies a constant digital value K to the adder 20. As described below, the value K determines the free running frequency of the DTO 30.

The signal provided by the adder 20 is applied to one input port of an adder 32 of the DTO 30. The output signal of the adder 32 is applied to the input port of an N-bit parallel-input parallel-output register 34. The output port of the register 34 is coupled to a second input port of the adder 32. The output port of the register 34 is the output port of the DTO 30. The clock input terminal of the register 34 is coupled to receive the signal, CK, provided by the oscillator 40. Synchronous with each pulse of the signal CK, the sum of the value held by the register 34 and the value provided by the adder 20 is stored into the register 34. Since the register 34 is an N-bit register, the largest value which it can hold is $2^N - 1$. When the sum, S, provided by the adder 32 is greater than $2^N - 1$, the value stored in the register 34 is S modulo 2^N , a value that is less than $2^N - 1$. Consequently, the value, S, provided by the register 34 will increase in value in a stepwise-linear manner over several clock periods and then decrease in value during one clock period. The signal, SW, represented by this changing value is a quantized sawtooth waveform. The frequency, f_{sw} , of the signal SW is described by the equation (1)

$$f_{sw} = ((K + FD)/2^N) f_{ck} \quad (1)$$

where f_{ck} is the frequency of the clock signal CK. The signal SW is applied to the address input port of a read-only memory (ROM) 36. The ROM 36 is programmed to provide output values representing a quantized sine wave signal in response to the values of the quantized sawtooth wave signal SW. The output signal provided by the ROM 36 is applied to a digital-to-analog converter (DAC) 38 which develops an analog signal approximating a sinusoidal signal having a frequency of f_{sw} .

The signal provided by the DAC 38 is applied to one input port of a phase comparator 52 of the reconstruction filter 50. A second input port of the phase comparator 52 is coupled to receive a signal, provided by the counter 60, which has a frequency substantially equal to f_{sw} . The phase comparator 52 produces an output signal which is proportional to the instantaneous difference in phase between the signals provided by the counter 60 and DAC 38. This output signal is applied to the input terminal of a loop filter 54 which integrates the phase difference signal to develop a signal which is proportional to the difference in frequency between the signals provided by the counter 60 and the DAC 38.

The frequency difference signal provided by the loop filter 54 is applied to the frequency control input terminal of an analog voltage controlled oscillator (VCO) 56. The VCO 56 is conditioned to produce a sinusoidal output signal having a frequency which is an integer multiple of the frequency f_{sw} . The signal produced by the VCO 56 is applied to wave shaping circuitry 58 which changes the sinusoidal signal to a square wave signal. The signal provided by the wave shaping circuitry 58 is the signal CLOCK which is applied to the ADC 12 and counter 60 as set forth above.

FIGURE 2 is a block diagram of a partly digital, partly analog phase locked loop system which includes the present invention. The circuitry shown in FIGURE 2 is analyzed in the context of an NTSC television signal processing system. The items which may be different for the circuitry to operate in a PAL television signal processing system are noted in parentheses where appropriate.

In FIGURE 2, a source 210 provides composite video signals to an ADC 212. The ADC 212 samples the signals provided by the source 210 using a sampling clock signal, CLOCK', which has a frequency substantially equal to, for example, 13.5 MHz. The signal CLOCK' is provided by the reconstruction filter 250 as set forth below. The digitized composite video signal developed by the ADC 212 is applied to synchronizing signal separating circuitry 214 which extracts the horizontal line synchronizing signal components from the digitized composite video signal and provides the extracted signal components as a signal HSYNC' to a phase comparator 216. The phase comparator 216 which, in this embodiment of the invention, is identical to the phase comparator 16 described in reference to FIGURE 1, provides an output signal representing the instantaneous difference in phase between the signal HSYNC' and a signal HS', having approximately the same frequency as the signal HSYNC'. The signal HS' is generated by a counter 260 which divides the frequency of the signal CLOCK' by 858 (864 for PAL).

The phase difference signal generated by the phase comparator 216 is applied to a loop filter 218. The filter 218 integrates the phase difference signal to generate a signal, FD', representing the difference in frequency between the signals HSYNC' and HS'. The signal FD' is summed by an adder 220 with a digital value K, provided by a digital value source 222. In the present embodiment of the invention, K is equal to

14,564. The signal developed by the adder 220 is applied to a DTO 230 which includes an adder 232 and a 16-bit parallel-input parallel-output register 234. The sample values provided by the adder 220 are summed with the value held in the register 234 by the adder 232, the resultant sum is loaded into the register 234 synchronous with the positive-going transitions of a 20 MHz clock signal CK' . The signal CK' is provided by a conventional resonant crystal controlled oscillator 240. The output signal, SW' , of the DTO 230 is the successive sample values provided by the register 234.

As set forth above, the signal provided by a DTO may be considered to be a quantized sawtooth signal having a frequency which depends on the frequency of the clock signal and the value of the increment signal applied to the DTO. In the present embodiment of the invention, when the signal FD' provided by the loop filter 218 is zero valued, the value in the register 234 will be incremented by the value $K = 14,564$ for each cycle of the 20 MHz clock signal CK' . Table I illustrates successive sample values of the signal SW' .

TABLE I

TIME	SW'
T_0	0
T_1	14,746
T_2	29,492
T_3	44,238
T_4	58,984
T_5	8,194
T_6	22,940
T_7	37,686
T_8	52,432
T_9	1,642
T_{10}	16,388
T_{11}	31,134
T_{12}	45,880
T_{13}	60,626
T_{14}	9,836
T_{15}	24,582

The values in Table I are shown graphically in FIGURE 3A. The times T_0 - T_{15} occur coincident with the positive going edges of the clock signal CK' and correspond to instants at which the value provided by the adder 232 are loaded into the register 234. In FIGURE 3A, the sawtooth waveform of the signal SW' is shown by the broken line connecting the sample points. In FIGURE 3A, the value P-1 represents the largest value which can be held by the register 234 (i.e. 65,535 or $2^{16}-1$). The value P/2 corresponds to 32,768 (i.e. 2^{15}).

When the signal FD' is nonzero, the value of the increment applied to the DTO is changed. The increment value is greater when FD' is greater than zero and less when FD' is less than zero. As set forth above, the frequency of the signal produced by the DTO is directly proportional to the increment value.

In a first embodiment of the invention to be described, a signal MSB_{sw} , representing the most significant bit (MSB) of the signal SW' is applied as a reference signal to the phase comparator 252 of the reconstruction filter 250. The connection applying the signal MSB_{sw} to the phase comparator 252 is illustrated by the alternating long and short dashed line 235. The signal MSB_{sw} is shown in FIGURE 3C. This signal has a value of logic zero when the sample values of the signal SW' are less than P/2 and a value of logic one when the sample values the signal SW' are greater than or equal to P/2. The signal MSB_{sw} may be considered to be the superposition of a signal having a frequency of f_{sw} , the frequency of the signal SW' , and of a beat signal having a frequency related to the harmonics and subharmonics of the respective signals SW' and CK' . The frequency of the beat signal component has a fundamental component that is substantially equal to the difference between the frequency, f_{sw} , and the subharmonic of the signal CK' which is closest in frequency to f_{sw} . In the present embodiment of the invention, f_{sw} is 4.5 MHz and f_{ck} , the frequency of the clock signal CK' , is 20 MHz. Accordingly, the beat signal has a fundamental frequency substantially equal to 500 KHz (5 MHz - 4.5 MHz). It is contemplated that other combinations of clock frequency and sawtooth wave frequency may be used as long as two limitations are observed: first, the ratio of the frequency of the clock signal to the frequency of the sawtooth wave should

be greater than two to satisfy the Nyquist criterion; second, this ratio should be a noninteger value having a fractional part that is large enough to ensure that the fundamental component of the beat frequency will be outside of the tracking range of the ADC. In general, the frequency of the fundamental component of the beat signal is largest when the fractional part of the ratio is one-half. As set forth below, the fundamental component of the beat signal is substantially attenuated by the reconstruction filter 250.

A phase comparator 252 suitable for use in this embodiment of the invention is described in U.S. Patent 4,520,319, entitled "Electronic Phase Detector Having An Output Which Is Proportional To The Phase Difference Between Two Data Signals", which is hereby incorporated by reference. The signal provided by this phase comparator has an average magnitude which is proportional to the difference in phase between the signal MSB_{sw} and a signal having a frequency approximately equal to f_{sw}' that is generated by the counter 260 in response to the signal $CLOCK'$. This phase difference signal is applied to a loop filter 254. The filter 254 integrates the phase difference signal to provide a signal which is proportional to the difference in frequency between the signal MSB_{sw} and the signal provided by the counter 260 to the phase comparator 252. This frequency difference signal is applied to a conventional analog VCO 256 which produces a sinusoidal output signal having a frequency of approximately 13.5 MHz. The output signal of the VCO 256 is applied to wave shaping circuitry 258 which converts the sinusoidal signal to a square wave signal. The square wave signal generated by the wave shaping circuitry 258 is applied to the counter 260 to complete the loop. The output signal of the wave shaping circuitry 258 is the output signal, $CLOCK'$, of the PLL system shown in FIGURE 2.

The counter 260 used in this embodiment of the invention is a cascade combination of two counters. The first counter (not shown) produces an output pulse for every three pulses of the signal $CLOCK'$. The output signal of this counter is applied to the phase comparator 252. The second counter (not shown) produces an output pulse for every 288 (288 for PAL) pulses of the signal provided by the first counter. The output signal of the second counter is the signal HS' which is applied to the phase comparator 216 as set forth above.

In this first embodiment of the invention, the reconstruction filter 250 acts to substantially attenuate the beat frequency components of the MSB of the signal SW' and to multiply the frequency of its 4.5 MHz signal component to obtain the 13.5 MHz signal, $CLOCK'$.

As set forth above, the frequency of the beat signal components of the signal MSB_{sw} depends on the relative frequencies of the harmonics and subharmonics of the signals SW' and CK' . In the embodiments of the invention illustrated in FIGURE 2, the frequency of the signal SW' changes to track changes in the frequency of the signal $HSYNC'$ provided by the synchronizing signal separation circuitry 214. It has been determined by the inventor that when the composite video signals provided by the source 210 are provided by a nonstandard source, such as a video tape recorder (VTR), the signal $HSYNC'$ may vary in frequency by as much as five percent from its nominal value. Consequently, the signals SW' and CK' may generate fundamental beat signal frequencies ranging between 275 KHz and 500 KHz.

The reconstruction filter 250 acts to attenuate these beat signal components because the passband of filter 250 encompasses only those frequencies which differ by approximately 15 KHz from a center frequency of three times the frequency of the signal SW' . The pass-bandwidth of the reconstruction filter 250 is determined by the loop time constant of the PLL which constitutes the filter 250. This time constant, which defines the frequency tracking range of the PLL, may be selected by adjusting the gain of the various elements in the PLL as described in an article by Wolf J. Gruen entitled "Theory of AFC Synchronization", Proceedings of the IRE, August 1953, pp. 1043-1048.

Since the fundamental frequency of the beat signal component of the signal MSB_{sw} is outside of pass-band of the reconstruction filter 250, the signal, $CLOCK'$, produced by the filter is relatively free of distortion related to this fundamental frequency component of the beat signal.

Distortion from the beat signal is not entirely absent from the signal $CLOCK'$, however. To understand the nature of this residual distortion, consider the frequency spectrum of the signal MSB_{sw} . This signal may be modeled as a square wave signal derived from a superposition of a first sinusoidal signal having a frequency of f_{sw} and a second signal which is not necessarily sinusoidal, having the frequency spectrum of the beat signal. It is well known that the frequency spectrum of a square wave includes the fundamental frequency and its odd harmonic frequencies. Consequently, the frequency spectrum of the signal represented by the signal MSB_{sw} includes components at the odd harmonic frequencies of each of the components of the beat signal as well as the odd harmonic components of the signal SW' . When the components of the beat signal have relatively large amplitude values and frequencies that lie within the pass-band of the reconstruction filter, they may cause significant distortion in the signal $CLOCK'$.

This distortion is substantially eliminated by circuitry delineated by the broken line block, 241. The circuitry 241 delays the negative and positive going edges of the signal MSB_{sw} by varying amounts of time

to substantially eliminate any distortion of the signal MSB_{sw} that is related to the beat signal.

In this embodiment of the invention, the amount of time by which a given transition of the signal MSB_{sw} is to be delayed is determined by examining five bits of the corresponding sample value of the signal SW , having less significance than the MSB. As set forth above, and as shown in FIGURE 3A, the sample values provided by the DTO 230 correspond to samples of a sawtooth waveform having a frequency of f_{sw} . Since the signal SW is a sawtooth wave signal having known minimum and maximum amplitude values, the timing of a given sample relative to the start of a period of the sampled waveform is proportional to the value of the sample. For example, in the example shown in FIGURE 3, the signal SW has a minimum value of zero and a maximum value of $P-1$. A sample at the midpoint of a period of the signal SW has a value of $P/2$. In the present example, the value P is 2^{16} (65,536). Accordingly, the sample value which corresponds to the midpoint of a period of the signal SW is 2^{15} (32,768). This sample value has a logic one in the 16th or MSB position and logic zeros in the 15 least significant bit (LSB) positions. In this example, therefore, the value of the 15 LSB's of a sample of the signal sw are indicative of the timing of the sample relative to the beginning or midpoint of a period of the signal SW when the MSB of the sample is logic zero or logic one, respectively. Consequently, the 15 LSB's of the sample value corresponding to a transition of the signal MSB_{sw} , or with adequate accuracy only the 5 MSB's of these 15 LSB's, can be used to adjust a programmable delay element to align the transition with the beginning or midpoint of a period of the signal SW .

Referring to FIGURE 2, the signal MSB_{sw} is applied to the input terminal of a conventional D-type flip-flop 236. The flip-flop 236 delays the signal MSB_{sw} by one period of the signal CK , applied to the flip-flop 236 by the oscillator 240. The signal MSB_{sw} and the delayed signal provided by the flip-flop 236 are applied to respectively different input terminals of an exclusive-OR gate 238. The gate 238 is a transition detector for the signal MSB_{sw} . It produces a logic one output signal only when the delayed MSB_{sw} signal provided by the flip-flop 236 differs from the undelayed signal MSB_{sw} provided by the DTO 230. The signal V_{sw} , the five MSB's of the 15 LSB's of the signal SW , is applied to the address input port of a ROM 244. The output port of the ROM 244 is coupled to the input port of a latch 246. The latch 246 is controlled to load a new value when the exclusive OR gate 238 has a value of logic one.

On the occurrence of a transition, the five-bit output value provided by the ROM 244 in response to the signal V_{sw} is loaded into the latch 246. The output port of the latch 246 is coupled to the control input port of a variable delay element 248. The signal input port of the delay element 248 is coupled to receive the delayed signal MSB_{sw} provided by the flip-flop 236 as further delayed by a compensating delay element 237. The delay element 237 delays the individual transitions in signal MSB_{sw} by an amount of time sufficient to match the transitions to the respective delay element control values produced by the ROM 244.

The variable delay element 248 used in this embodiment of the invention is a chain of 31 pairs of inverters (not shown). The output terminal of each pair of inverters is coupled to a respectively different input port of a multiplexer (not shown). The five-bit control signal provided by the ROM 235 is coupled to the control input port of the multiplexer to select either the input signal to the delay element 248 or the output signal provided by one of the 31 pairs of inverters as the output signal, MSB_{sw} of the variable delay element 248. A variable delay element suitable for use as the delay element 248 is described in U.S. Patent No. 4,489,342 entitled "Mosfet Integrated Delay Circuit For Digital Signals and Its Use in Color Television Receivers".

The amount of time by which the variable delay element 248 delays a transition of the signal MSB_{sw} is determined by the values provided by the ROM 244 via the latch 246. These values, in turn, are determined by the values of the signal V_{sw} , representing bits 11-15 of the signal SW . As set forth above, the values of the signal V_{sw} , which correspond to transitions in the signal MSB_{sw} , are proportional to the timing of the transition relative to the beginning or midpoint of the sawtooth waveform depending on whether the transition is positive-going or negative-going, respectively. In the present embodiment of the invention, it is desirable to delay the positive-going transitions of the signal MSB_{sw} so that they occur substantially coincident with the endpoint or positive peak of the sawtooth waveform and to delay the negative going transitions of the signal MSB_{sw} so that they occur substantially coincident with the midpoint of the sawtooth waveforms.

For example, as shown in FIGURES 3A through 3D, it is desirable to delay the transitions of the signal MSB_{sw} occurring at times T_3 , T_5 and T_7 by the respective amounts of time d_1 , d_2 and d_3 to produce the transitions in the waveform MSB_{sw} occurring at times T'_3 , T'_5 and T'_7 .

The amount of time by which an individual transition is to be delayed is inversely proportional to the corresponding value of the signal V_{sw} . In this embodiment of the invention, the ROM 244 is programmed to translate the values of V_{sw} into delay control values. First, the range of values of the signal V_{sw} is divided into 32 equal parts and then 32 delay control values which span one-half of a period of the signal sw are

assigned to the cells of the ROM 244 that are addressed by the respective values of the signal V_{sw} . An example of the programming of the ROM 244 is presented below in Table II. In this example, the ROM 244 has 32 addressable storage locations numbered from zero to 31 which subdivide the values represented by the 15 LSB's of the signal SW' into 32 equal parts represented by their 5 MSB's.

TABLE II

Address	Delay Value	Address	Delay Value	Address	Delay Value
0	31	13	18	26	5
1	30	14	17	27	4
2	29	15	16	28	3
3	28	16	15	29	2
4	27	17	14	30	1
5	26	18	13	31	0
6	25	19	12		
7	24	20	11		
8	23	21	10		
9	22	22	9		
10	21	23	8		
11	20	24	7		
12	19	25	6		

When a value of zero is selected from the ROM 244 for application to the control input port of the variable delay element 248, the output signal, MSB'_{sw} is the same as the input signal to the delay element 248. When a value between 1 and 31 is selected, however, the output signal MSB'_{sw} is equal to the signal MSB_{sw} , applied to the variable delay element 248, delayed by a chain of a number of inverter pairs equal to the selected value.

The circuitry 241 generates a signal MSB'_{sw} having transitions which are aligned with the midpoints and endpoints of the sawtooth waveform of the signal SW' . This alignment is accurate to within the time delay defined by one inverter pair of the variable delay element 248. Assuming that f_{sw} is 4.5 MHz and that the inverter pairs of the variable delay element 248 subdivide one-half of the period of the signal SW' into 32 equal time intervals, each of the transitions of the signal MSB'_{sw} is accurate to within 3 ns relative to a square wave signal having a frequency of f_{sw} .

The signal MSB'_{sw} produced by the variable delay element 248 is applied to the reconstruction filter 250 in place of the signal MSB_{sw} applied via the connection 235. The PLL of the reconstruction filter 250 is synchronized by the signal MSB'_{sw} as set forth above to produce the signal $CLOCK'$, which has a frequency substantially equal to 13.5 MHz and which is substantially free of any distortion related to the beat signal components of the signal MSB_{sw} .

Claims

1. Apparatus comprising:

a variable oscillator (256) responsive to a control signal for generating an oscillatory signal having a frequency determined by the value of said control signal;

digital signal generation means (230) including a digital oscillator for generating a plural-bit sampled data digital oscillatory signal (SW') having a frequency which approximates a predetermined nominal frequency; characterized by:

digital phase comparison means (253) coupled to said variable oscillator and to said digital signal generation means for generating a phase difference signal proportional to the difference in phase between said oscillatory signal and a single-bit digital signal (MSB'_{sw}) representing a predetermined bit of said plural-bit digital oscillatory signal; and

means (254) coupled to said digital phase comparison means and responsive to said phase difference signal for generating said control signal.

2. The apparatus set forth in Claim 1 wherein:

said digital signal generation means includes a source (240) of clock signal (CK') having a predetermined frequency; and

said digital oscillator (230) is responsive to said clock signal for generating samples of said plural-bit sampled data digital oscillatory signal at instants determined by said clock signal; and characterized in that:

the ratio of the frequency of said clock signal to the nominal frequency of said plural-bit sampled data digital oscillatory signal is a noninteger value greater than two.

3. The apparatus set forth in Claim 2 including:

means (241) coupled to said digital oscillator for generating said signal representing said predetermined bit of said plural-bit sampled data digital oscillatory signal, said generating means including means (244-248) for delaying the occurrence of transitions (236,238) in the value of the predetermined bit (MSB_{sw}) of said sampled data digital oscillatory signal by respective amounts of time that are inversely proportional to the respective values represented by bits (V_{sw}') having less significance than said predetermined bit, of respective samples of said sampled data digital oscillatory signal, and corresponding to the transitions in the value of said predetermined bit.

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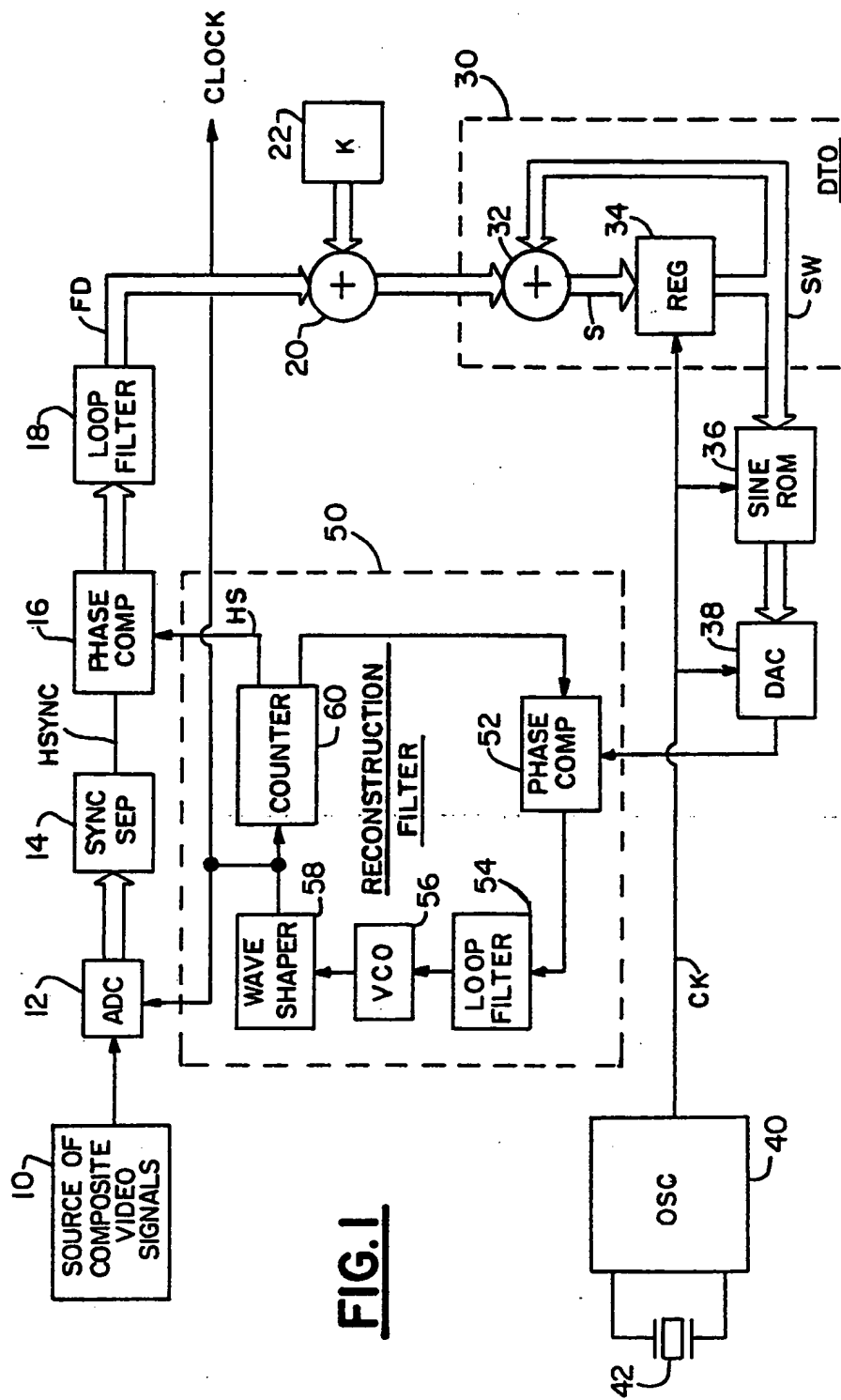
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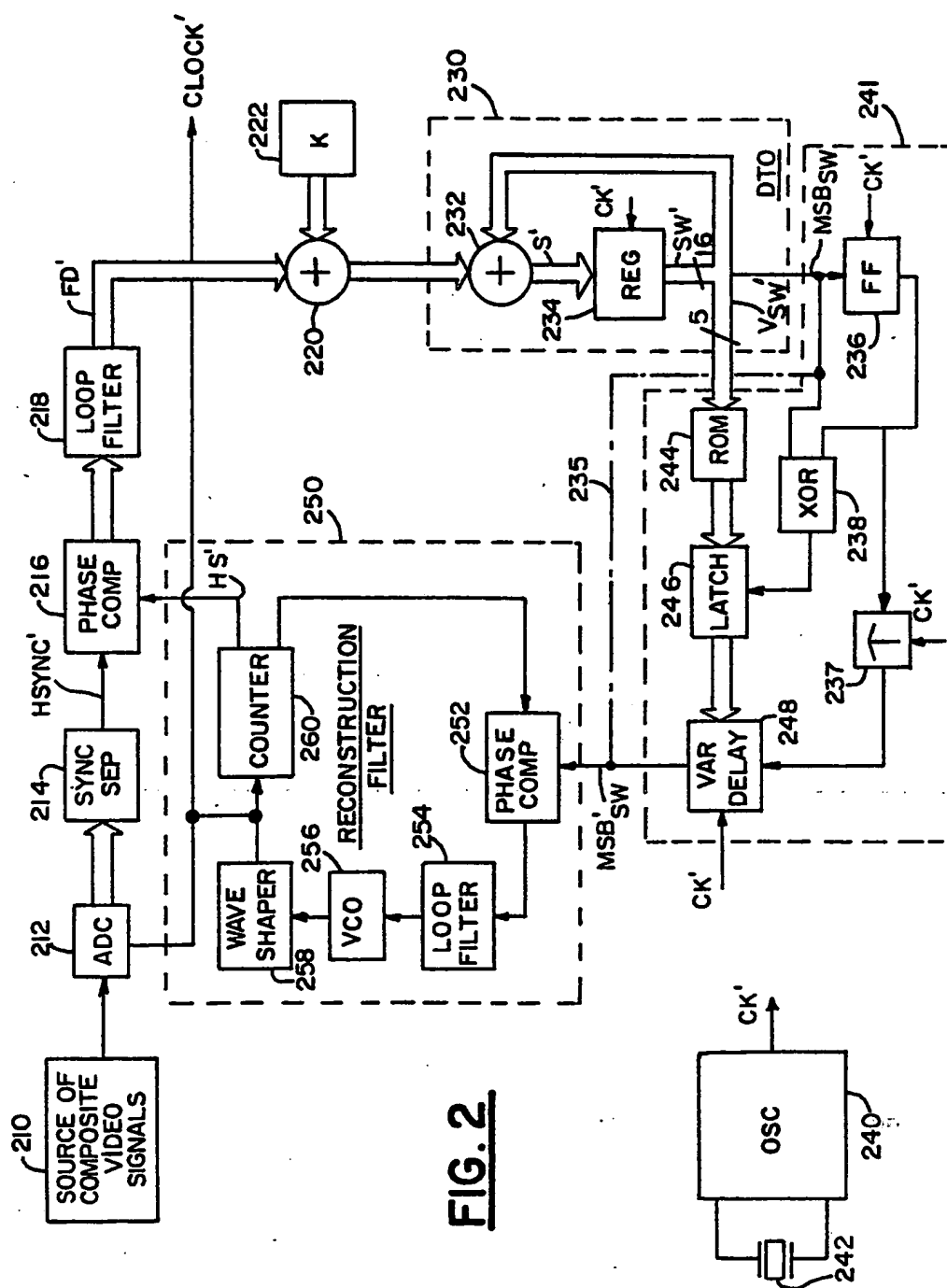
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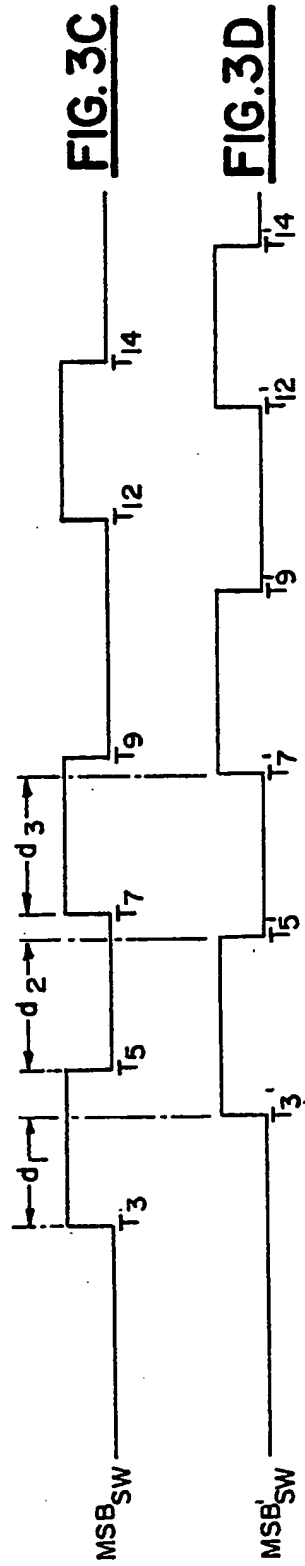
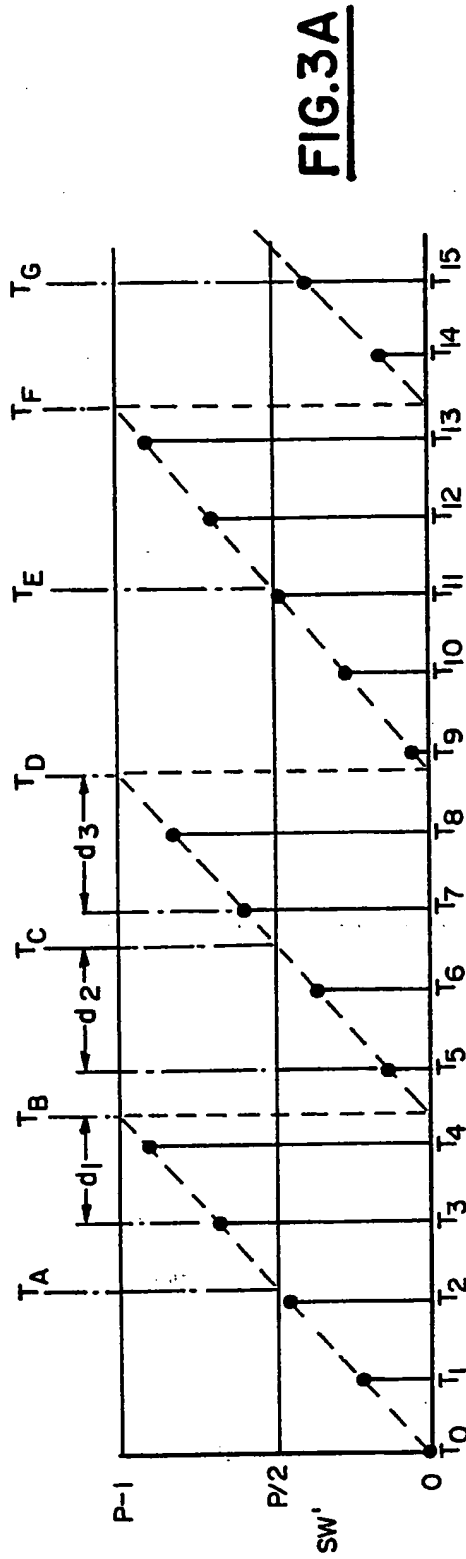
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PRIOR ART





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